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Transmitted herewith for filing is the patent application of

Inventor(s): ALAIN GEORGES

For (Title): INTERACTIVE DIGITAL MUSIC RECORDER AND PLAYER

Enclosed are.

☒ 10 sheets of informal drawings.☐ An assignment of the invention to _____☒ A declaration and power of attorney (signed).☒ A verified statement to establish small entity status under 37 CFR 1.9 and 37 CFR 1.27.☐

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APPLICATION TRANSMITTAL

☒ Applicant Alain GEORGES ☐ Patentee
☒ Application No. 09/ ☐ Patent No.
☒ Filed on 10/17/2000 ☐ Issued on

Title: INTERACTIVE DIGITAL MUSIC RECORDER AND PLAYER

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16-OCT-2000

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Date

INTERACTIVE DIGITAL MUSIC RECORDER AND PLAYER

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Background of the Invention

(a) Field of the Invention

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The current invention relates to a device that provides for interaction with a user during recording, playing, composing and modifying musical selections.

(b) Description of Related Art

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Various devices capable of composing a musical piece automatically have been developed, such as that described in US Patent number 4,399,731. These devices are based on synthesizers of instrumental sounds on which are applied, for instance as per the MIDI (Musical Instrument Digital Interface) standard, "scores" of notes that are composed automatically.

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Various devices capable of recording or playing video recordings have already been developed, such as digital or analog camcorders and other video recorders. These devices can possess one or several audio inputs, which permit mixing or replacement of the sound track that was recorded originally during the recording of the image with an external audio source.

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Furthermore, there also exist various devices that are capable of playing recorded pieces, such as digital compact disc players, players of compressed files (for instance as per the MPEG-level 3 standard), etc. Finally, there exist devices incorporating a tuner, which permit reception of radio broadcasts via electromagnetic waves.

Summary of the Invention

An interactive digital multi-media device provides a user with multiple related features, wherein it is not necessary that the user have training in musical or sound handling arts. A programmable memory is used to store digital audio and voice samples. External sound provides external sound signals, such as radio and TV signals that may also be stored. Playback means is provided connected to the stored audio and voice samples and the external sound signals. Automatic musical composition instructions are stored so that the user is able to create unique musical compositions. Components are provided that function to mix externally obtained sound with the unique musical compositions and stored audio and voice samples to provide altered audio and voice presentations.

The present invention includes an improved automatic composition device that adds to instrumental sounds some previously recorded sound samples wherein the sound samples are mixed with a melody in the process of the automatic composition. The sound samples will typically, but not exclusively, be human voice samples.

The invention herein further represents an improvement over prior art devices by integrating in a video recorder or player a module that generates music or other sounds which either can be mixed with the originally recorded sound (sound mixing), or can replace the originally recorded sound (sound dubbing). This sound mixing or dubbing can be performed either at video/audio record time or at play time.

Additionally the invention envisions selection of musical pieces either out of a library stored for instance in the form of digitized musical files, MIDI files or other types of files, or from pieces that are composed in a pseudo-random fashion using, for example, a synthesizer function to play original musical pieces.

The invention further envisions selection of musical pieces according to a specified musical style, in a pseudo-random fashion, or according to predefined criteria, wherein the audio files meet the predefined criteria and are either extracted from a music library or are generated by an automatic composition function. The resulting sounds are used either during the video recording or during the video playback.

The present invention represents an improvement of the aforementioned devices by simulating the reception of a radio station by the use of one or several of the aforementioned

functions. Further, the invention permits selection of musical pieces, by a user, either from a library in which musical pieces are stored as compressed musical files, MIDI files or other similar types of files, from pieces recorded from the output of a radio receiver, or from pieces that are composed in a pseudo-random fashion using a synthesizer function to play original musical pieces.

Thus, the present invention provides for selection according to a pre-selected musical style, in a pseudo-random fashion or according to a pre-defined criteria, of audio files to be played from a speaker, wherein the audio files meet the pre-defined criteria and are either extracted from the library or generated by an automatic composition function. Further, the recording of or the automatic generation of sentences that mimic the speech of a “disc-jockey” or of an announcer permits combination of speech passages with the musical pieces being played, thereby giving the user the illusion that he is listening to an actual radio station.

Brief Description of the Drawings

The above objects and other advantages of the present invention will become more apparent by describing in detail the preferred embodiments of the present invention with

5 reference to the attached drawings in which:

Figure 1 is a perspective of the digital multi-media device of the present invention.

Figure 2 is a functional block diagram of the present invention.

Figure 3 is a block diagram of the automatic composer of the present invention.

10 Figure 4 is a block diagram showing one structure for summing outputs in the present invention.

Figure 5 is a block diagram showing another structure for summing outputs in the present invention.

Figure 6 is a block diagram showing a first variant of the diagram of Figure 3.

Figure 7 is a block diagram showing a second variant of the diagram of Figure 3.

15 Figure 8 is a block diagram of another embodiment of the automatic composer of the present invention.

Figure 9 is a block diagram of a variation of the present invention.

Figure 10 is a block diagram of another aspect of the present invention.

Figure 11 is a block diagram of an additional aspect of the present invention.

20 Figure 12 is a block diagram of yet another aspect of the present invention.

Detailed Description of the Preferred Embodiments

The present invention will be described in greater detail with reference to certain preferred and alternative embodiments. As described below, refinements and substitutions of the various embodiments are possible based on the principles and teachings herein.

The invention disclosed herein is an interactive digital music player that allows one or more users to listen to, compose, and interact with music in any environment. The device is a hand-held digital music player that offers numerous unique features that, until now, have not existed in a single music product. One of the many features, the Electronic DJ (e-DJ) automatically composes and plays in real-time music in a multitude of genres (such as a dance or techno), which can be mixed with user-supplied voice samples. The e-DJ offers a unique interactive way of playing with music. At any moment, the user can take the control of the music played by the e-DJ via an attractive interface (joystick, graphical display). The user can change the music patterns played by various instruments, change the relative level, apply effects, play pre-recorded samples, etc.

Further, the disclosed device allows the user to listen to, create, download, store, and interact with music, and includes an FM radio receiver. Users can listen to both compressed audio and MIDI karaoke music files, store music on a plug-in SmartMedia memory card (SSFDC storage device), and carry the unit to any location for playing. The device is capable of storing more than 1000 MIDI karaoke-file songs on a 64MB smart media memory card and it can provide over 120 minutes of digital music play time. In addition, the disclosed device is able to create music files, accept music files created by the user, download music from the Internet via a PC, take a music transferred from a PC, or, accept music added from any other smart media memory card. When connected to a docking station, the device offers additional features like insertion of Karaoke lyrics in a video source for display on a TV screen, MIDI PC connection or remote control.

Figure 1 shows the device 10 and its docking station 11. A number of cables 12 are shown for connection to a number of ports (not shown) situated at the rear of the docking station 11. The device 10 as shown has a group of controls 13, including buttons, knobs, jacks, etc. and a display 14 on its front surface. A connector within a slot 16 is shown on the device that is configured to mate with a connector (not shown) on a rear surface of a base 17 on the docking

station 11. The docking station includes an inclined surface 18 that supports the device in the docking station when connection is made between the connector in slot 16 and the connector in the docking station.

The device 10 is shown in the block diagram of Figure 2 and will be referred to as a Digital Multi Media Platform (DMM) hereinafter. A large number of features are included in the DMM, one of which is the ability to playback digital audio data from a SmartMedia memory card 19. The digital audio source is either compressed or MIDI. The DMM offers a user the capability to interact with a song being played by changing pitch and tempo. The audio compression techniques supported in the DMM is Microsoft Windows Media Audio (WMA), illustrated at 22 in Figure 2, having files in the active streaming format (ASF). WMA provides a higher compression ratio for equivalent sound quality as compared to previous compression techniques, i.e., MP3. DMA compression techniques allow a compression ratio approximately twice that of MP3, thereby allowing for over two hours of music to be stored on a 64 megabyte SmartMedia card. The WMA playback function supports the Digital Rights Management (DRM) technology that is used with audio files. This technology can be used to encrypt the compressed data, which can then only be decoded correctly with a private key. As a result, a user can download music from the Internet that is protected with the DRM technology and receive rights to play music on a particular DMM that has been previously registered. ASF and MIDI files can be copied on the SmartMedia card either via a serial PC connection offered by the DMM device or directly with an off the shelf SmartMedia card reader. Internal memory and SmartMedia memory 19 further provides the ability for the DMM to obtain firmware and sound upgrades from the Internet.

An electronic DJ and virtual radio functions are provided by a group of musical algorithms 21 that allow a user to automatically create and play many different types of music such as dance, techno, rap, easy listening, etc. Unlimited hours of random music may be generated by the musical algorithms. The device also operates as a musical composer assistant. An FM receiver 23 is provided in the DMM so that a user is able to tune to a favorite radio station or to a virtual radio station. While music is playing a user is able to take control over automatic composition being undertaken by the musical algorithms 21 and to add the users own touches. The user may interact with the music being played by way of changing the tempo and pitch. Consequently, interaction is available in as much as the user is able to play voice, music

or sound samples that can be recorded with a built in microphone or obtained from the FM radio receiver or any other source. The samples are obtained through the use of a high quality audio compression circuit 24 and are stored in the SmartMedia memory section 19. The controls 13 on the face of the DMM (Figure 1) provide for introducing the sound effects such as wobbler, dobbler, etc. and can be applied at a user's will during playback of the recorded samples.

Additionally the controls allow the user to take even more control over the musical content by selecting relative volume of an instrument or a group of instruments in a fashion normally only possible through the use of a mixing table. The sound of a particular instrument may be altered using filters and also may be saved to be later played over again and even to be edited later.

Controls 13 are intuitive using traditional player keys while the graphical display of 14 provides visual feedback in real time of the action the user is undertaking with the musical content. As a result users with no special musical education or skill are able to alter and compose music to their own designs.

Audio compression circuit 24 preferably is a circuit based on a speech compression algorithm using, for example, adaptive differential pulse coded modulation (ADPCM) that can be applied to a useful portion of the recording. As is known in the art, techniques such as PCM and ADPCM provide methods of compression of analog data that has been sampled and digitized, typically in which the samples have been digitized, and difference between samples are stored/sent (rather than the samples themselves), or with adaptive techniques a predicted sample/sample difference is computed, and what is stored/transmitted is the difference between the predicted value and the actual value, etc. Such compression techniques are known in the art and used, for example, in various telephony-type systems.

A built in microphone (not shown) in the DMM is used for Karaoke and DJ functions. Further, it is used with a music synthesizer 26 to provide for audio mixing and other audio sound effects. The synthesizer 26 provides a MIDI interface whereby, due to the small size of a MIDI file, a large number of songs may be made to fit in the flash memory 19 containing the SmartMedia cards. The SmartMedia Flash memory provides for storage of compressed digital/audio MIDI/Karaoke files and voice samples. This is currently the most compact Flash Memory form available. The digital nature of the MIDI files allows tempo and pitch of music to be dynamically adapted to the users requirements. Moreover, when a singer's voice is taken from the built in microphone within the DMM and mixed with synthesized or stored sound

samples, lyrics may be displayed on a TV/video interface 27 or on the display 14 of Figure 1. An external microphone may be plugged into the DMM for even more flexible usage.

When the DMM is connected to the docking station 11 and playing MIDI or Karaoke songs, a digital light show is available through the TV/video interface 27. The resulting colorful patterns have movements and shapes following the beat of specific instruments or serve as a compliment to lyrics displayed on the screen 14 or just as a visual assist to the electronic DJ music.

A universal serial connection 28 to a personal computer allows transfer of audio files to and from the SmartMedia/flash memory 19. Additionally the serial connection is available to perform a firmware upgrade. The MIDI interface 26 in the DMM 10 converts to a synthesizer when connected to a MIDI sequencer through the USB PC interface 28. Firmware for the DMM is stored in the Flash Memory and can therefore be upgraded readily by a user by downloading through the MIDI PC interface 28. A digital instrument tuner for all modern and classical instruments is shown in Figure 2 at 32.

The compact design obtained in the DMM to afford the portable nature of the device is obtained by housing some functions required only when using the product in a nonmobile environment to be placed in the docking station 11 of Figure 1. These functions include the circuitry required for the insertion of Karaoke lyrics in a video source and for display on a TV screen as well as an external microphone connection for Karaoke recording. A MIDI in/out interface is provided in the docking station 11 for connection to a personal computer or to another MIDI device. A power jack and charge control (not shown) is provided at the rear of the docking station to provide the DMM with a 3.3 volt power supply in a preferred embodiment as shown at 33 in Figure 2. The DMM is configured to include two AA batteries. These may be rechargeable or regular batteries. Multiple functions are provided through controls 13 on the DMM such as compressed audio playback, electronic DJ, voice and sound samples and radio. Playback and record keys are available on the face of the DMM together with general volume control keys. Power, video and audio connectors, and microphone input connectors are provided on the docking station 11. Additionally, the docking station has a standard MIDI interface with in and out connectors. This interface is useable for communication with a personal computer for audio file transfer or firmware upgrade.

The invention disclosed herein is useable by persons having no appreciable musical expertise. It presents a new way of listening to music wherein a user may modify existing music, compose new music, and save and play back music. Music patterns composed by the user or taken from favorite composers may be retained in a music database or library. The composition device of the present invention utilizes a combination of notes rather than sounds. Notes, being represented digitally, require less memory for storage than sounds that are usually represented in analog form and take large amounts of memory, comparatively, when digitized.

Further, digital notes are easily altered while digital sounds are not readily changed. As a result, the instant invention may use a microprocessor operating at about 25 MHz and needs only about two (2) megabytes of storage, although, of course, the particular clock speed and storage size are exemplary embodiments and the invention is not limited thereto. This may be seen to be much less processor speed and memory than used in the least capable desktop computer, for example.

According to the functional diagram of Figure 3, one typical embodiment of the automatic composition device of the present invention includes essentially a processor 36, a memory 37 containing a music database for use in the automatic composition device, a memory 38 for storing sound samples utilized by the automatic composition device, a music synthesizer 39, and a summation and digital to analog conversion circuit 41 interconnected as shown. Processor 36 is able to compose automatically a melody out of the database stored in memory 37 using an automatic composition algorithm containing composition instructions. The automatic composition algorithm also utilizes the sound samples stored in memory 38, in such a way that processor 1 delivers in synchronism on its output a control signal M1 connected to synthesizer 39 and a sound sample signal S2. An output signal M2 of the synthesizer and sound sample signal S2 are then summed and converted to analog form in circuit 41 to deliver the complete audio signal MA3. This audio signal MA3, which forms the output of the invention, can then be presented in audio form by a stereo sound system.

In a preferred embodiment of the invention, the processor is made of a microprocessor or microcontroller having a synchronizing clock and being linked to one or several memories. Clock speed and memory capacity may be relatively low, 25 MHz and 2 MB being ample in the preferred embodiments. A RAM memory (volatile memory) can serve as the working memory of the microprocessor and can be used to store the sound samples in memory 38, whereas a ROM

or EPROM memory can store the microprocessor program and the music database in memory 37 used by the automatic composition device. However, a greater flexibility will be granted by non-volatile memories: RAM memory saved by a disposable or rechargeable battery, or Flash EEPROM memory (electrically erasable). The non-volatile memory can be used at the very least to store the sound samples in memory 38, so that they are saved when the invention is powered off. It can also be used to store the music database in memory 37 for use by the automatic composition device, as well as a microprocessor operating program. This permits easy update of the music database and the microprocessor operating program by means of downloading updates.

The summing of output signal M2 of the synthesizer and of sound sample signal S2 is obtained using various methods and circuitry described as follows.

In a first scheme for performing the summing of M2 and S2 (Figure 4), output signal M2 of synthesizer 39, which is a digital signal, is converted to an audio signal MA2 by digital to analog converter 42. Sound sample signal S2 generated by processor 36, which is also a digital signal, is separately converted to an audio signal SA2 by a second digital to analog converter 43. Then audio signals MA2 and SA2 are summed by analog adder 44 to deliver an audio output signal MA3. The analog adder 44 can consist of an operational amplifier mounted in an adder configuration.

In a second manner of performing the summing of M2 and S2 (Figure 5), output signal M2 of synthesizer 39 and sound sample signals S2 generated by processor 36 are added by digital adder 46 to form a combined digital signal M3. This digital signal is then converted to an audio signal MA3 by a digital to analog converter 47. While this manner of performing the summing requires only one digital to analog converter 47, it nevertheless requires a digital adder 46.

In a variant of the second manner of performing the summing of M2 and S2 (Figure 6), output M2 of synthesizer 39 is sent back to processor 36, which then performs by itself the summing of digital signals M2 and S2 (in this variant, signal S2 only exists in a virtual form in the processor). Thus, the processor delivers directly on output M3 the compound digital signal, which is converted to an audio signal MA3 by digital to analog converter 47. This variant requires only one digital to analog converter and does not require a specific digital adder circuit, but conversely it requires an additional input on the processor to read the output of the synthesizer.

In a second variant of the second manner of performing the summing of M2 and S2 (Figure 7), which is applicable to the case where synthesizer 39 itself comprises a processor (microprocessor and/or digital signal processor), the summing of digital signals M2 and S2 can be performed by the synthesizer modified for this purpose. Indeed, either the synthesizer has an additional input to read signal S2 as shown in Figure 7, or signals M1 and S2 are multiplexed on a single input of the synthesizer (variant not shown on a figure). In the latter case, if a MIDI standard protocol is used, the multiplexing of signals M1 and S2 will be made much easier by putting the information relevant to sound sample signals S2 in "System Exclusive" MIDI messages. Thus, the synthesizer performs internally the summing of signals M2 and S2 (in this variant, signal M2 only exists in a virtual form in the synthesizer) and delivers directly on output M3 the compound digital signal, which is converted to an audio signal MA3 by digital to analog converter 47. This second variant also requires only one digital to analog converter and does not require a specific digital adder circuit, but conversely, compared with a standard synthesizer circuit, it requires a modification of the synthesizer to read and add sound sample signals S2.

In another preferred embodiment of the invention, applicable to the case where synthesizer 39 itself comprises a processor, memory 38 for storing sound samples is linked to synthesizer 39 instead of processor 36. Given that synthesizer 39 already possesses a memory that contains instrumental sounds that are used as basic units for music synthesis, a variant of this embodiment consists in storing the two types of sounds, i.e., the basic instrumental sounds and the sound samples, in a sounds memory 48 as shown in Figure 8. In this embodiment, the summing of digital signals M2 and S2 is naturally performed by the synthesizer, according to the same process as described for the variant of Figure 7. Thus, the program executed by the processor of the synthesizer 39 is modified to permit these new operations (storing sound samples and summing them with instrumental sounds). The program modifications performed by the processor of the synthesizer can be minimized by organizing the program in such a way that the synthesizer processes sound samples as a special case of basic instrumental sounds; for example, as by assigning a MIDI channel to the sound samples or as an alternative by defining the sound samples as special notes of a percussion type instrument.

In a preferred embodiment of the invention, the invention allows a user to record sound samples with a microphone via commands that permit the user to start and stop a recording. An internal microphone and/or an external microphone input is required. As seen in Figure 9, a

microphone 49 provides an output S1A1 connected to the input of preamplifier 51. An analog to digital converter 52 converts the microphone signal S1A1 to digital sample signals SI2. The sample signals are read by processor 36 and stored either in its memory 38, or, in the case of the embodiment of Figures 7 and 8, in the memory contained within the synthesizer 39. The memory may be RAM or flash as indicated previously.

Various processes may optionally be used to reduce the memory requirements to store the samples. The following examples are not limiting:

(a) once the recording is over, the silent periods which precede and follow the useful portion of the recording can be eliminated automatically;

(b) a speech compression algorithm such as Adaptive Differential Pulse Coded Modulation (ADPCM) 24 can then be applied to the useful portion of the recording.

In the process of automatic composition of music, the processor can treat the sound sample signals as a special case of a musical instrument. Thus, the association of sound sample signals with the notes sent by the instruments is an integral part of the automatic composition process. The processor then extracts information from the available pool of sound samples in the same way as from the music database.

However, the processing of sound samples has certain features. First of all, the invention is not restricted to using sound samples that have been pre-recorded for a given type of instrument, but also envisions using sound samples that have been recorded freely by the user. In addition to some special effects that will be described later, the processing of sound samples consists of selecting a sample in the pool of available samples according to a pseudo-random sequence, then selecting the moment to start playing the sample, taking into account the melody that the processor is simultaneously composing, in such a way that the sample starts at the beginning of a musical bar, or in specific cases at another moment that is governed by certain musical laws (which may depend on the music style). The sound sample can then be played either in its entirety or in part, and it can be repeated or not repeated.

Furthermore, the sound samples may be modified to include various special effects depending on the style of music created by the automatic composition device. To quote only a few of such special effects: echo addition, vibrato, distortion, frequency modulation, various filtering processes to shape the sound spectrum, etc.

One aspect of the current invention relates to a system such as described above that also may include, or be included in, an automatic soundtrack generator that operates to merge an independent sound track with a video sequence during recording or playback of the sequence (aspects of the other embodiments described herein could be used to record or playback desired audio tracks in conjunction with video that being recorded and/or played back, etc.). Figure 10 represents a system in block diagram form showing an external video input 53 providing a signal V1 to a video and sound recording module 56 connected to store the signal V1 on storage unit 58 such as a tape or a digital memory. The storage unit is connected to a video and sound playback module 61 that produces an output V2 connected to a video output module 62 and an audio output A2 connected to a sound generation module 57. Items 56, 58, 61 and 62 are generally contained in a commonly known camcorder device. The camcorder device also includes an external sound input element 54, such as a microphone, producing an audio output A1. Output A1 is also connected to sound generation module 57. A control module 59, shown connected to items 56, 58, 61 and 57 in Figure 10, selects sounds generated by sound generation module 57 to produce output S2 providing audio output 63.

According to the functional diagram of Figure 11, the invention can typically be embodied by a module that generates music or other sounds and which comprises essentially processor 36, memory 37 containing the music database of the automatic composition device, musical synthesizer 39, memory 38 storing the sound samples for the musical synthesizer, a digital to analog conversion circuit 43, a summation circuit 58 and a memory 42, internal or external to the invention, containing a library of digitized musical files. The memory elements 37, 38 and 42 can be made of one or several distinct physical components. Processor 36 is able to select, according to certain criteria that are pre-defined or defined by a user, musical files out of the library in memory 42, or is able to compose automatically a melody out of the database stored in memory 37, with an automatic composition algorithm. The output signal S3 of the synthesizer, after digital to analog conversion in converter 43, delivers an analog signal A3 which can optionally be mixed with the recorded audio A2 or with the external audio input A1 in summation circuit 58 to deliver the mixed complete audio signal S1/S2. This mixed audio signal S1/S2, which forms the output of the invention, can then be used as a sound source at video/audio record time (real time) (S1) or at play back time (time shifted) (S2). As a result

prerecording of movie soundtracks, for example, is unnecessary. A simple way is provided for a user to change sound content, pitch, etc. for implementation in a video soundtrack.

In a preferred embodiment of the invention, the processor 36 is made of a microprocessor or microcontroller linked to one or several memories. A RAM memory (volatile memory) can serve as the working memory of the microprocessor, whereas a ROM or EPROM memory can store the microprocessor program and the music database 37 of the automatic composition device. However, a greater flexibility will be granted by non-volatile memories: RAM memory saved by a disposable or rechargeable battery, or Flash EEPROM memory (electrically erasable). The non-volatile memory can be used to store the music database 37 of the automatic composition device and the digitized musical files of the library in sound samples memory 38, as well as the microprocessor program. This permits easy update for the music database and the microprocessor program.

The synthesizer 39 and the microprocessor 36 of Figure 11 cooperate to select banks of sound samples according to predetermined processor instructions to provide the synthesizer output S3. The sound sample banks contain sounds in digital form of predetermined instruments.

The processor 36 can also be used to select the sound source or sources described above. Furthermore, the processor can be used to select commands which permit activation functions such as playing a recording, selecting and mixing an audio source, or controlling any other functions that are commonly found in video recorder based equipment. This combines the functions of blocks 57 and 59 of Figure 10.

According to another feature of the present invention, a device is provided that simulates a radio station including a player of musical pieces, either recorded and digitized or synthesized. According to the functional diagram of Figure 12, a typical embodiment of the invention comprises essentially a processor 36, a memory 37 containing a music database for use by an automatic composition algorithm, a memory 38 storing the sound samples, a musical synthesizer 39, a summation and digital to analog conversion circuit 43, a radio receiver 64 and a memory 42, internal or external to the invention, containing a library of digitized musical files, wherein these elements are interconnected as shown. The memory elements 37, 38 and 42 can be made of one component or several physically distinct components. Processor 36 is in communication with the memory elements and is able to select, according to certain criteria, musical files out of the library of musical files or is able to compose automatically, according to the automatic

composition algorithm, a melody out of the database stored in memory 37. The automatic composition algorithm also utilizes the sound samples stored in memory 38, which may include some speech sentences, in such a way that processor 36 delivers in synchronism on its outputs a control signal M1 connected to synthesizer 39 and a sound sample control signal S2. Output signal M2 of the synthesizer and sound sample control signal S2 are then summed and converted to analog form in circuit 43 that provides the complete audio signal MA3 for connection to a speaker or speakers (not shown). In a similar way, the output of radio receiver 64 can be mixed upstream, as a digital signal, or downstream, as an analog signal, of circuit 43 to add a supplementary sound source to the complete audio signal MA3. The audio signal MA3 forms the output of the invention that can then be played by the aforementioned speakers in a stereo system.

In a preferred embodiment of the invention, the processor is made of a microprocessor or microcontroller linked to one or several memories. A RAM memory (volatile memory) can serve as the working memory of the microprocessor and can be used to store the sound samples 38, whereas a ROM or EPROM memory can store the microprocessor program and the music database 37 used by the automatic composition algorithm. However, a greater flexibility will be granted by non-volatile memories: RAM memory saved by a disposable or rechargeable battery, or Flash EEPROM memory (electrically erasable). The non-volatile memory can be used at the very least to store the sound samples in memory 38, so that they are saved when the invention is powered off. It can also be used to store the music database in memory 37 for use by the automatic composition algorithm, the digitized musical files of the library 42, as well as a microprocessor operating program. This permits easy update of the music database and the microprocessor operating program by means of downloading updates.

Processor 36 can also be used to select the sound source or sources, including radio receiver 64 or one of the sources in memories 37, 38 and 42 described above. It can also be used to select one of a number of preset radio stations, including either actual radio stations or illusory radio stations. Illusory stations in reality play sound samples and musical pieces that have been stored in digital form or that are composed automatically as described herein. Moreover, a sound input device, such as a microphone 66, is useful to input voice signals, encoded in memory, to be used in construction of illusory radio station sound patterns.

The controls 13 to which reference was made in Figure 1 include a joystick and a number of direct access keys to simplify entry into various interactive modes by users of the DMM device of the present invention. Certain operating sequences for specific interactive modes will now be described.

- 5 To enter Electronic-DJ, press the [e-DJ] direct access key. The LCD will display "E-DJ" in the status line and a selection of music styles is proposed: Techno, Dance, Hip-Hop, etc.

To select the desired style, use Joystick Up/Down

- To start playing auto-composed music of the selected style, press the play/pause key. the LCD will display the music I-way screen representing six instrument lanes: Drums, Bass, Riff,
10 Lead, Samples and Microphone. The screen is animated with sound waves or pulses synchronized with music beats.

To change music tempo, press Pitch/Tempo key together with joystick left to speed up or right to slow down.

- To change music pitch, press Pitch/Tempo key together with joystick left to speed up or
15 right to slow down.

To change music pitch, press Pitch/Tempo key together with joystick up to higher the pitch or down to lower it.

Press play/pause to pause the music.

Press stop to stop the music and go back to style selection screen.

- 20 Press forward to start a new song.

Press backward to restart the current song.

Press Save/Edit key to save the current song as a user song on the SmartMedia card.

To select a specific lane, use joystick left/right.

The image on the screen will shift left or right and present the I-way seen from the new lane.

- 25 To change the relative volume of the current lane, press Effects key together with the joystick up or down.

To change the reverb on the current lane, press Effects key together with the joystick left or right.

- To enter the Underground mode for the current lane and loop the current sequence, press
30 joystick down.

The transition to music tunnel is marked by a screen animation and the under ground mode is entered for the selected instrument or sound source (sample or microphone). The U-ground screen shows the inside of a tunnel with sound waves on both sides and beat pulses coming towards the user. The shape of the far side of the tunnel is modified with the sound effect applied.

The music will loop as long as the user remains in the Underground mode.

In instrument tunnels (Drums, Bass, Riff and Lead):

Use Joystick right to select a new music pattern. Use joystick left to come back to previous patterns.

Press Joystick up to go back to music I-way.

To change the cutoff frequency of the filter on the current instrument, press Effects key together with joystick up or down.

To change the resonance of the filter on the current instrument, press Effects key together with joystick left or right.

Press stop to mute the instrument.

Press play/pause to un-mute the instrument.

In the sample tunnel:

Use Joystick left/right to select a sample.

Press Joystick up to go back to music I-way.

To change the sample volume, press Effects key together with joystick up or down.

To choose a different sample effect, press Effects key together with joystick left or right.

Press play/pause to play the sample. If sample playback had been previously disabled (see below), the first press on play/pause will re-enable it. Following presses will play the selected sample.

Pressing stop will disable the automatic playback of samples by the e-DJ when returning to I-way mode.

In the microphone tunnel:

Use Joystick left/right to select the active microphone: built-in or docking station. If the docking station is not connected, no selection is available.

Press Joystick up to go back to music I-way.

To change pitch on the microphone input, press Pitch/Tempo key together with joystick up or down.

To change the microphone volume, press Effects key together with joystick up or down.

To modify the microphone echo, press Effects key together with joystick left or right.

5 Press stop to mute microphone.

Press play/pause to un-mute the microphone.

To exit Electronic-DJ mode, choose any other mode by pressing the appropriate direct access key.

10 Playback of compressed audio and Karaoke is accessible in the “Songs” mode. This mode allows to play digitally recorded WMA (MP3) songs, MIDI songs, Karaoke songs and User songs saved during an e-DJ session.

To enter Songs mode, press the [Songs] direct access key. The LCD will display “e.Songs” in the status line and a list of available songs or song lists on the SmartMedia card to choose from. Song lists are identified by a specific icon.

15 To select the desired song or list, use Joystick Up/Down.

To start the playback of selected song or song list, press the play/pause key. The LCD will display the play song screen. The name of the song is scrolling in a banner in the center right part of the LCD while the audio output level is materialized by a frame around the name which size is changing following the audio level. The status line shows the elapsed time.

20 If the song is a Karaoke song, the Lyrics are displayed at the bottom of the LCD. Note: if the song is a user song (composed with the e-DJ), the music I-way mode is entered instead of the play song mode.

To change music tempo, press Pitch/Tempo key together with joystick left to speed up or right to slow down.

25 To change music pitch, press Pitch/Tempo key together with joystick up to higher the pitch or down to lower it.

Press play/pause to pause the music. Press play/pause again to resume playback.

Press stop to stop the music and go back to song selection screen.

Press forward to go to next song.

30 Press backward to go to previous song.

Pressing the Save/Edit key in the song select screen or while a song is playing enters the song edit mode. Depending on the type of the song (user song, MIDI or WMA), different parameters can be edited.

Pressing the Save/Edit key in the song select screen while the current item is a song list enters the song list edit screen. See "Editing items" below.

Pressing forward key in the song select screen will create a new song list.

To exit Songs mode, choose any other mode by pressing the appropriate direct access key.

Playback of compressed audio and Karaoke is accessible in the "Songs" mode. This mode allows to play digitally recorded WMA (MPA3) songs, MIDI songs, Karaoke songs and User songs saved during an e-DJ session.

To enter Songs mode, press the [Songs] direct access key. The LCD will display "e.Songs" in the status line and a list of available songs or song lists on the SmartMedia card to choose from. Song lists are identified by a specific icon.

To select the desired song or list, use Joystick Up/Down.

To start the playback of selected song or song list, press the play/pause key. The LCD will display the play song screen. The name of the song is scrolling in a banner in the center right part of the LCD while the audio output level is materialized by a frame around the name which size is changing following the audio level. The status line shows the elapsed time.

If the song is Karaoke song, the Lyrics are displayed at the bottom of the LCD. Note: if the song is a user song (composed with the e-DJ), the music I-way mode is entered instead of the play song mode.

To change music tempo, press Pitch/Tempo key together with joystick left to speed up or right to slow down.

To change music pitch, press Pitch/Tempo key together with joystick up to higher the pitch or down to lower it.

Press play/pause to pause the music. Press play/pause again to resume playback.

Press stop to stop the music and go back to song selection screen.

Press forward to go to next song.

Press backward to go to previous song.

Pressing the Save/Edit key in the song select screen or while a song is playing enters the song edit mode. Depending on the type of the song (user song, MIDI or WMA), different parameters can be edited.

Pressing the Save/Edit key in the song select screen while the current item is a song list enters the song list edit screen. See "Editing items" below.

Pressing forward key in the song select screen will create a new song list.

To exit Songs mode, choose any other mode by pressing the appropriate direct access key.

In the "playing samples" mode, the user can record or play voice, music or sound samples.

To enter Samples mode, press the [Samples] direct access key. The LCD will display "e.Samples" in the status line and a list of available samples or sample lists on the SmartMedia card to choose from. Sample lists are identified by a specific icon.

To select the desired sample or list, use Joystick Up/Down.

To select the desired sound effect, press Effect key together with joystick left/right.

To change the sample volume, press Effect key together with joystick up/down.

To start the playback of the selected sample, press the play/pause key. The LCD will display the play sample screen. The name of the sample is scrolling in a banner in the center right part of the LCD while the audio output level is materialized by a frame around the name which size is changing following the audio level. The status line shows the current effect.

Press stop to stop the sample and go back to sample selection screen.

Pressing the Save/Edit key in the sample select screen or while a sample is playing enters the sample edit mode.

Pressing the Save/Edit key in the sample select screen while the current item is a sample list enters the sample list edit screen. See "Editing items" below.

Pressing forward key in the sample select screen will create a new sample list.

To exit Sample mode, choose any other mode by pressing the appropriate direct access key.

Recording samples is a simple operation possible in almost any operating mode of the dB1.

Press record and keep record button down to record a sample. Release the record button to end the recording. Recording is stopped automatically if the sample duration exceeds 30 seconds.

The record source is chosen automatically depending on the operation mode.

5 While e-DJ is playing (I-way or U-ground modes), the record is inactive.

If no music is playing, the record source is the active microphone (built-in or docking station).

If music is playing (songs or radio), the record source is a mix of the music and the microphone input if not muted.

10 To enter Virtual radio mode, press the [v-Radio] direct access key. The LCD will display "v.Radio" in the status line and a list of available station presets to choose from as well as the currently tuned frequency. If no preset has been stored, only the currently tuned frequency is displayed.

To select the desired station, use Joystick Up/Down.

15 To listen to the selected station, press the play/pause key. The LCD will display the radio screen. The name of the radio station (or frequency if it is not a stored preset) is scrolling in a banner in the center right part of the LCD. An animation representing radio waves is also displayed. The status line shows the tuned frequency.

Use Joystick left/right to go to previous/next station in the preset list.

20 Press play/pause to mute the radio.

Press stop to go back to station preset selection screen.

Press forward to tune up.

Press backward to tune down.

25 Pressing the Save/Edit key in the station preset selection screen enters the preset edit mode.

Pressing the Save/Edit key while a station is playing will store it in the preset list.

To exit v.Radio mode, choose any other mode by pressing the appropriate direct access key.

30 As long as no music is playing and in the v.Radio mode, the microphone settings are made as follows:

To change the pitch on the microphone input, press Pitch/Tempo key together with joystick up to increase the pitch or down to lower it.

To change the microphone volume, press Effects key together with joystick up/down.

To modify the microphone echo, press Effects key together with joystick left/right.

5 Reference is also made to co-pending U.S. App. Ser. Nos. _____ and _____ filed on even date herewith for "Virtual Radio" and "Automatic Soundtrack Generator," both of which are incorporated by reference herein and both of which may utilized, alternatively, with embodiments of the present invention.

10 Although the invention has been described in conjunction with specific preferred and other embodiments, it is evident that many substitutions, alternatives and variations will be apparent to those skilled in the art in light of the foregoing description. Accordingly, the invention is intended to embrace all of the alternatives and variations that fall within the spirit and scope of the appended claims. For example, it should be understood that, in accordance with the various alternative embodiments described herein, various systems, and uses and methods
15 based on such systems, may be obtained. The various refinements and alternative and additional features also described may be combined to provide additional advantageous combinations and the like in accordance with the present invention. Also as will be understood by those skilled in the art based on the foregoing description, various aspects of the preferred embodiments may be used in various subcombinations to achieve at least certain of the benefits and attributes
20 described herein, and such subcombinations also are within the scope of the present invention. All such refinements, enhancements and further uses of the present invention are within the scope of the present invention.

What is claimed is:

1. An interactive digital music device providing multiple features for a user,
comprising
5 programmable memory for storing digital audio and voice samples,
receivers for obtaining external sound signals,
means for playing said stored audio and voice samples and external sound signals,
means for storing automatic musical composition instructions, whereby the user is able to
create unique musical compositions, and
10 means for mixing external sound signals with said unique musical compositions and
stored audio and voice samples to provide altered audio and voice presentations.
2. The interactive digital music device of claim 1, comprising
means for providing visual effects complementing said altered audio and voice
presentations.
- 15 3. The interactive digital music device of claim 2, wherein said means for providing
visual effects, comprises
visual means for providing written song lyrics.
4. The interactive digital music device of claim 2, wherein said means for providing
visual effects, comprises
20 visual means for providing complimentary light patterns for said altered audio and voice
presentations.
5. The interactive digital music device of claim 1, wherein said programmable
memory comprises

flash memory, and wherein said received radio signals, unique musical compositions and altered audio and voice presentations are digitized.

6. The interactive digital music device of claim 1, comprising
a television signal interface.

5 7. An Automatic Composer in a digital multi-media device for composing a musical piece according to automatic composition instructions and for mixing sound samples into it, comprising

a processor having an input and an output, said processor operating to implement the automatic composition instructions,

10 a memory for storing a music database for use in accordance with the automatic composition instructions,

a second memory for storing the sound samples, said first and second memories being connected to the processor input, a music synthesizer connected to the processor output for control by said processor and providing a synthesizer output, and

15 a summation and digital to analog conversion circuit for receiving processor and synthesizer outputs and for providing a summed analog output.

8. The Automatic Composer as set forth in claim 7, wherein said summation and digital to analog conversion circuit, comprises

20 a digital to analog converter for receiving said synthesizer and processor outputs and for providing analog signals, and second a summation circuit for receiving said analog signals.

9. The Automatic Composer as set forth in claim 7, wherein said summation and digital to analog circuit comprises a digital adder for summing the processor and synthesizer outputs and for providing a summed digital output, and a digital to analog converter for receiving said summed digital output.

25 10. The Automatic Composer as set forth in claim 9, wherein said processor comprises an internal digital adder.

11. The Automatic Composer as set forth in claim 9, wherein said synthesizer comprises an internal digital adder.

12. The Automatic Composer as set forth in claim 7, wherein said synthesizer comprises an additional processor for directly accessing said memory for storing sound samples, and wherein said synthesizer comprises means for handling sound samples as a special case of basic instrumental sounds.

13. The Automatic Composer as set forth in claim 7, comprising a microphone for recording external sound samples, and means for starting and stopping recording.

14. The Automatic Composer as set forth in claim 13, further comprising means for automatically eliminating the silent periods that precede and follow a useful portion of a recording, and means for implementing a speech compression algorithm to compress the useful portion of the recording.

15. The Automatic Composer as set forth in claim 7, wherein said processor comprises

means for integrating said sound samples into musical compositions,

means for functioning to select said sound samples according to a pseudo-random sequence,

means for directing said sample to be played at a predetermined time between the beginning and the end of a musical bar as governed by certain musical rules,

means for directing any portion of the sound sample to be played from its entirety to any part thereof, and

means for optionally selecting repetition of the sound sample.

16. The Automatic Composer as set forth in claim 7, wherein said processor comprises means for imparting special effects to the musical piece, whereby the musical piece is optionally modified with any ones of echo, vibrato, distortion, frequency modulation, and filtering effects.

17. The Automatic Composer as set forth in claim 7, wherein said processor comprises

a clock operating at 25 MHz maximum, and wherein said first memory and said second memory comprise

a memory having a capacity of 2 MB maximum.

18. An Automatic Soundtrack Generator in a digital multi-media device that is capable of mixing a signal from an external audio source in a video/audio record and/or play device with an internal audio source containing digitized sound files in accordance with predetermined mixing instructions, comprising

5 a processor,
a memory element connected to said processor for storing mixing the instructions and the digitized sound files,
a digital to analog conversion circuit for receiving the digitized sound files, and
a summation circuit for combining the signal from the external audio source with the
10 digitized sound files.

19. The Automatic Soundtrack Generator as set forth in claim 18, further comprising
a digital musical synthesizer, and
a store of pre-recorded and automatically composed MIDI files, wherein selected ones of
said store of MIDI files are connected to said digital musical synthesizer for providing the
15 internal audio source.

20. The Automatic Soundtrack Generator as in claim 18, further comprising
a user accessible input, and
means for selecting the internal audio source connected to the user accessible input and
defined by the user.

21. The Automatic Soundtrack Generator as defined in claim 18, further comprising
an additional memory means containing preprogrammed instructions for defining the
internal audio source, and
means for selecting the internal audio source connected to said additional memory means.

22. The Automatic Soundtrack Generator as set forth in claim 18, comprising
25 means for selecting the internal audio source at video/audio record time.

23. The Automatic Soundtrack Generator as set forth in claim 18, comprising
means for selecting the internal audio source when playing a previously recorded
video/audio sequence.

24. A Virtual Radio in a digital multi-media device that is capable of selecting,
30 according to given criteria, musical files from predetermined sound sources for playing from a
speaker, comprising

a processor providing a digital output,
a memory element connected to said processor for storing the given criteria and the
predetermined sources of musical files, and

a summation and digital to analog conversion circuit connected to receive said digital
5 output and providing a predetermined analog sound output.

25. The Virtual Radio device as set forth in claim 24, further comprising
a digital musical synthesizer for adding MIDI files that are pre-recorded or composed
automatically, according to the given criteria, to the predetermined sound sources.

26. The Virtual Radio device as set forth in claim 24, comprising
10 a radio receiver providing a sound output used as an additional selectable sound source.

27. The Virtual Radio device as set forth in claim 26, comprising means for providing
voice files within the predetermined sound sources for combination with said radio receiver
sound output.

28. The Virtual Radio device as set forth in claim 24, wherein sound samples are pre-
15 recorded in the predetermined sound sources by a user, comprising means for mixing speech
sound with the musical files.

[illegible]

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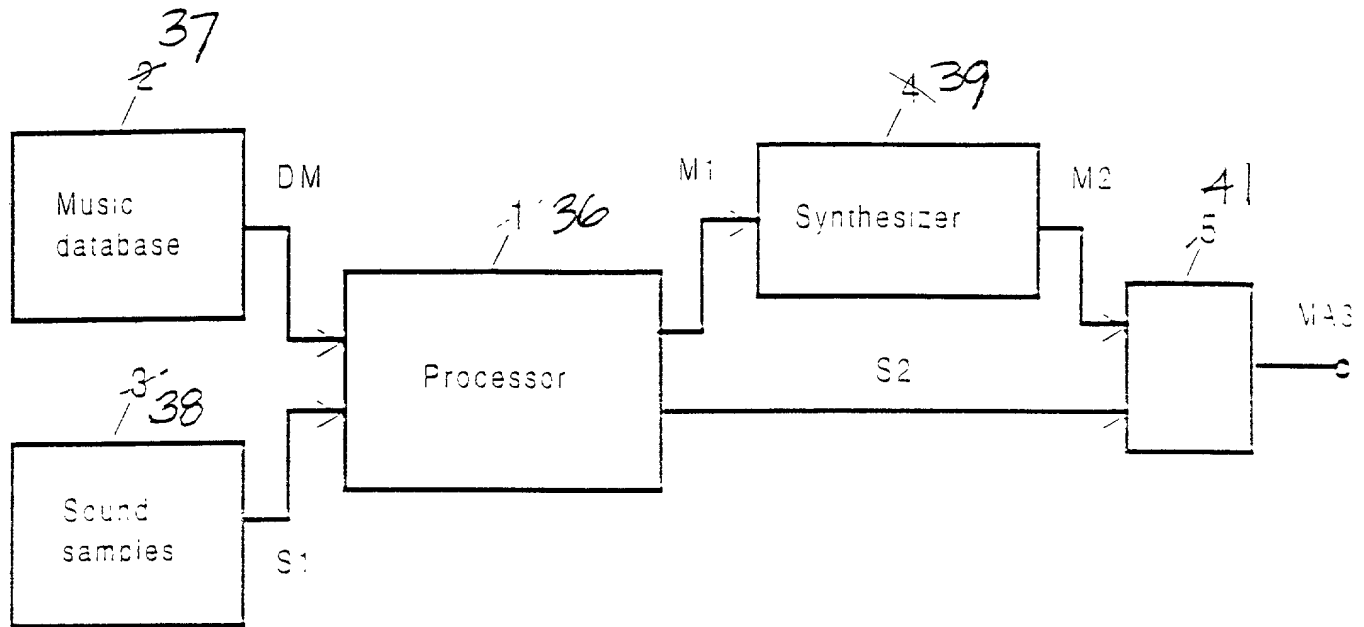


FIG. 3

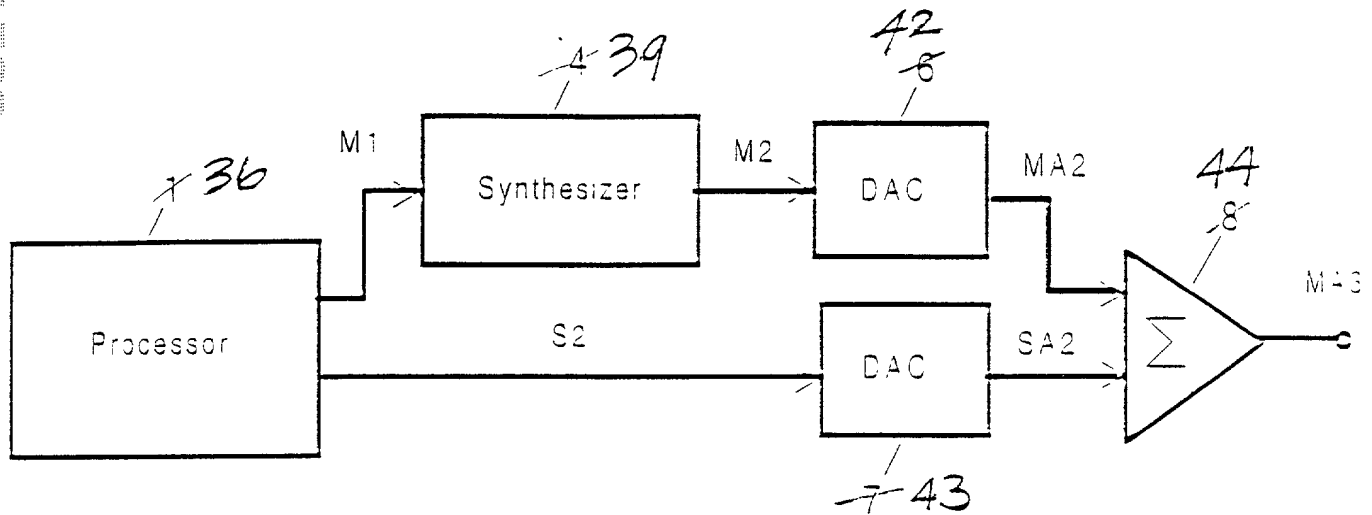


FIG. 4

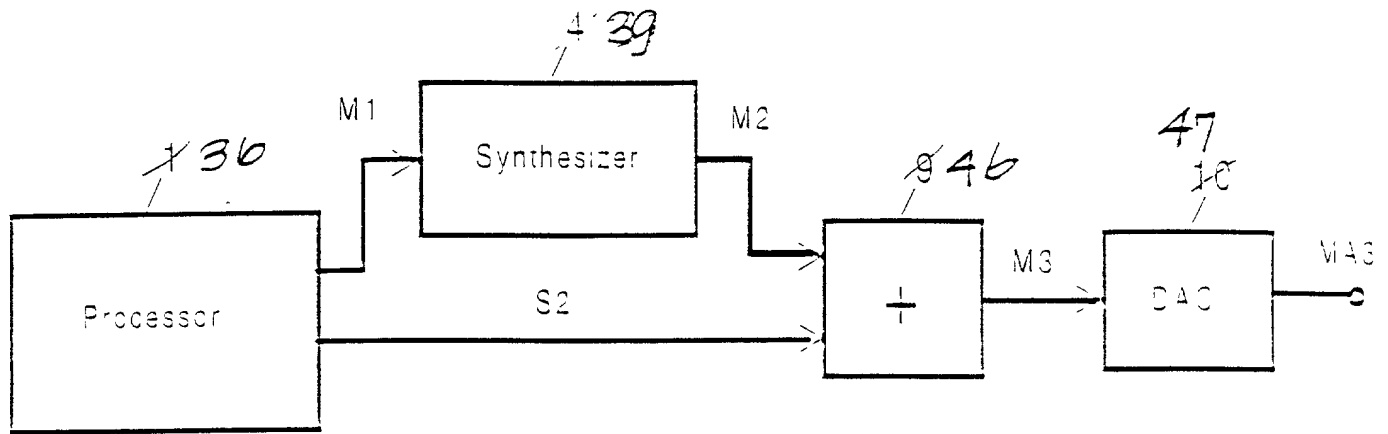


FIG. 5

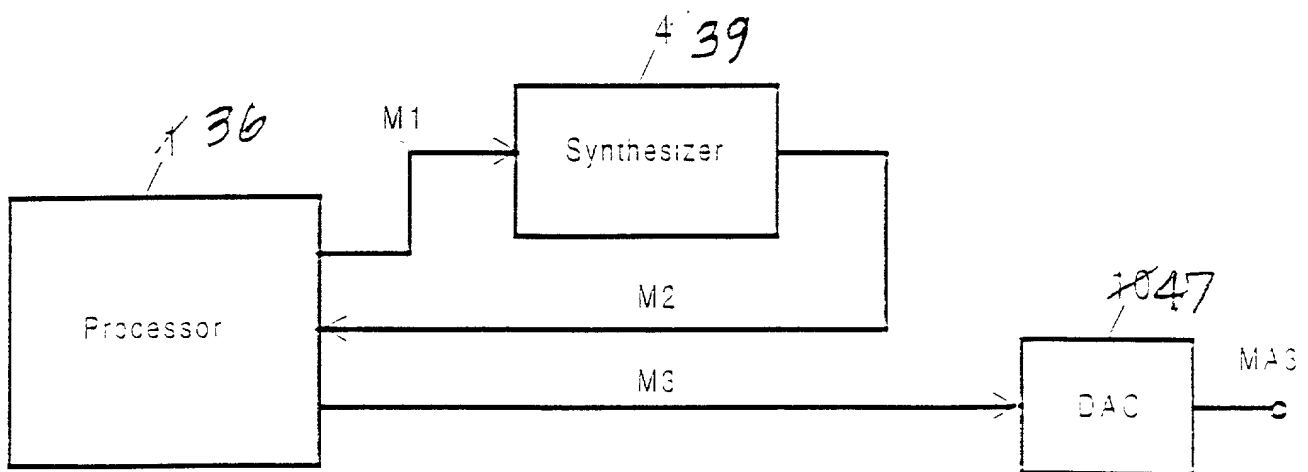
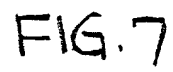


FIG. 6

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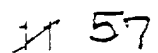


FIG. 10

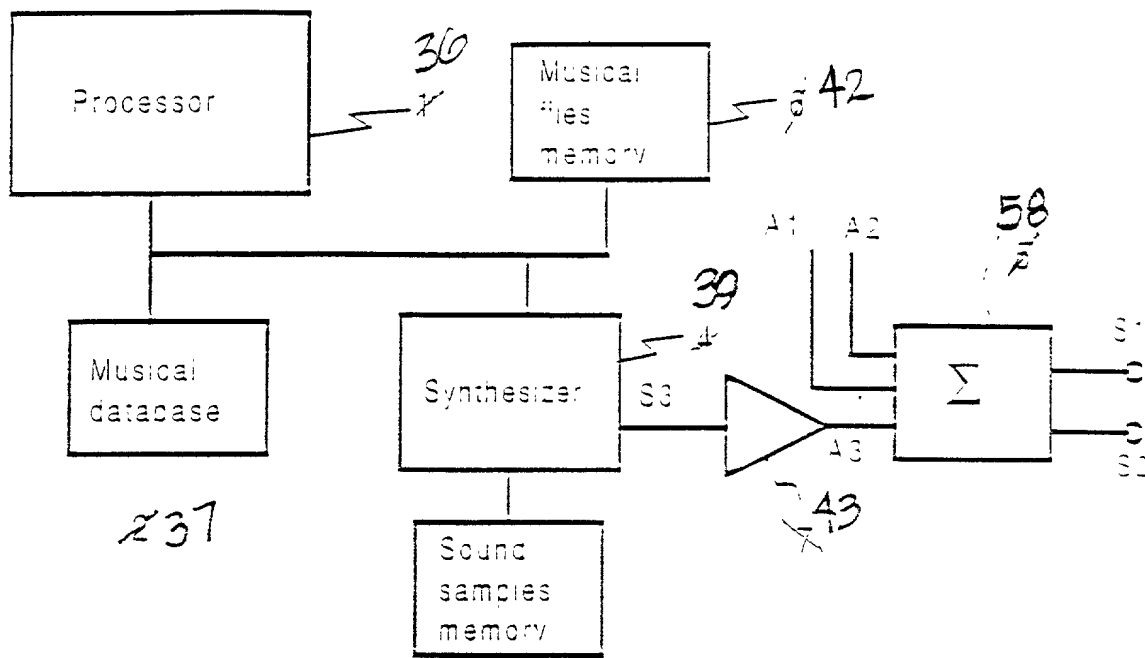


FIG. 11

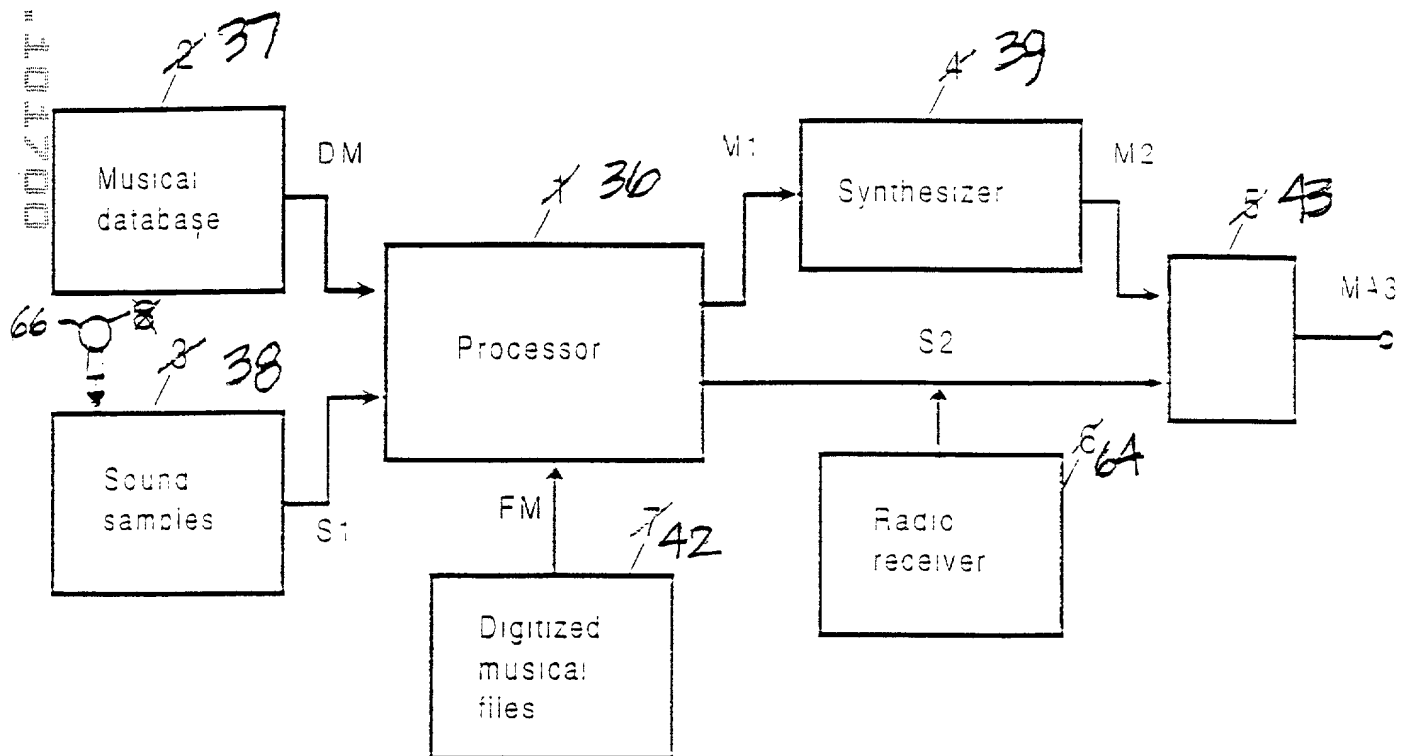


FIG. 12

DECLARATION AND POWER OF ATTORNEY

As a below named inventor, I hereby declare that:

INVENTOR AND SPECIFICATION IDENTIFICATION

My residence, post office address and citizenship are as stated below next to my name, I believe that I am the original, first and sole inventor (*if only one name is listed below*) or an original, first and joint inventor (*if plural names are listed below*) of the subject matter which is claimed and for which a patent is sought on the invention entitled:

INTERACTIVE DIGITAL MUSIC RECORDER AND PLAYER

TITLE OF INVENTION

the specification of which:

☒ is attached hereto.

— was filed on _____ as Application Serial No. _____
and was amended on _____ (*if applicable*).

— was described and claimed in PCT International Application No. _____ filed on _____
and amended under PCT Article 19 on _____ (*if any*).

REVIEW OF PAPERS AND ACKNOWLEDGMENT OF DUTY OF CANDOR

I hereby state that I have reviewed and understand the contents of the above identified specification, including the claims, as amended by any amendment referred to above.

I do not know and do not believe that the invention claimed in the above-identified specification was ever known or used in the United States of America before my or our invention thereof, or patented or described in any printed publication in any country before my or our invention thereof or more than one year prior to this application, and that the same was not in public use or on sale in the United States of America more than one year prior to this application.

I acknowledge the duty to disclose to the Patent and Trademark Office information which I know is material to the patentability of this application in accordance with Title 37, Code of Federal Regulations, § 1.56.

FOREIGN APPLICATIONS AND PRIORITY CLAIM

The invention claimed in the above-described specification has not been patented or made the subject of an inventor's certificate issued before the date of this application in any country foreign to the United States of America on an application filed by me or my legal representatives or assigns more than twelve months prior to this application. I hereby claim foreign priority benefits under Title 35, United States Code, § 119 of any foreign application(s) for patent or inventor's certificate or of any PCT international application(s) designating at least one country other than the United States of America listed below and have also identified below any foreign application(s) for patent or inventor's certificate or any PCT international application(s) designating at least one country other than the United States of America filed by me on the same subject matter having a filing date before that of the application(s) of which priority is claimed.

fine or imprisonment, or both, under Section 1001 of Title 18 of the United States Code, and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

HAND PRINT DATE BEFORE SIGNING

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Inventor's signature _____ Date _____

Residence _____

Post Office Address _____

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